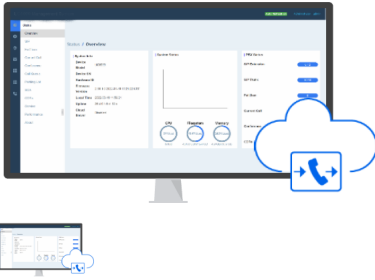


TS8000



TS8000 VoIP PBX

?? SIP/IP PBX(Software-based SIP/IP PBX)

?: ???
1-4 Weeks
★ ★ ★ ★ ★

[????????](#)

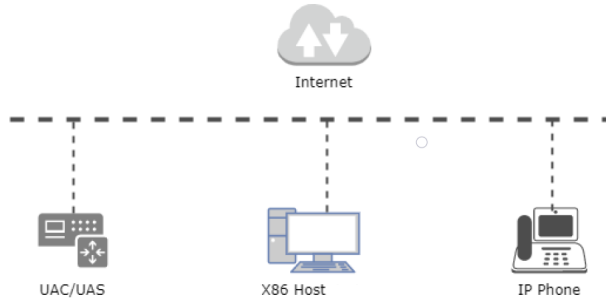
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TS8000????????SIP/IPPBX
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? TS8000????????SIP/PSTN/IMS????????????????????????????????IVR??

TS8000????20,000?SIP???4,000????????TS8000?????IPPBX?UC?SIP
?????? TS8000????????????????????????????????IP????????????????

TS8000????????????????????X86?ARM???????????? TS8000
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???????? TS8000
????????
????????????????????IPPBX????????????TS8000
????????????????????Dell/IBM????????????PSTN???TONE
SPREAD????TS2500??

TS8000????????????SIP/IP
????????????????????IPPBX????????????SIP/IMS????????????VoIP
????????????????????VoIP???



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- ?? 20,000 ? SIP ????? 4,000 ?????
- ?????X86?ARM?Docker?AWS?Google????
- ??TONE SPREAD????TS2500
- ?????????????/????????????
- ?????????????????????
- ??IVR????IVR????/????
- ?????????????????????

IP PBX Features

3-way Calling, Conference call
 Video call
 Voicemail, Voicemail to Email
 Voicemail Forwarding
 Call Control
 Call with Password Protection
 Call Priority
 Call Group Control
 Instant meeting, Scheduling meeting(Audio-only)
 Blacklist/Whitelist
 CDRs/Call Signaling Recording
 One Touch Recording
 Auto-recording
 Playback recording on web
 One SIP account with multi device registrations
 One device Multiple Numbers
 Auto Provisioning
 Auto-attendant Function
 Multi-level IVRS
 Feature codes
 Designated Pickup
 Caller ID display
 Manager/Secretary Function
 Caller/Called Number Manipulation
 Routing Based on Time Period
 Routing Based on Caller/Called Prefixes
 Attendant Console
 Mobile Extension
 Auto-Configuration
 IP Blacklist
 Extension User Management Interface
 Multi-language System Prompt
 Random Password for Extension

Media Capabilities

VoIP Protocols: SIP over UDP/TCP, RFC3261
 SDP, RTP/RTCP, SSL
 Audio Codecs: G.711a/u, G.723, G.729, G.722, G.726, Opus
 Video Codecs: VP8, H261, H263, H264, H263-1998, H263-2000
 Voice Activity Detection(VAD)
 Comfort Noise Generator(CNG)
 Adaptive Dynamic Buffer
 Adjustable Gain Control
 FAK, T.38 and Pass-through
 NAT, STUN/DNS
 DTMF: RFC2833/SignalInband
Call Features
 Call Forward (Always/No Answer/Busy)
 Call Forwarding for particular user
 Blind/Attended Transfer
 Redial/Call Return
 Speed dial
 Call transfer, Call parking, Call waiting
 Do-not-disturb (DND)
 DISA
 Music on Hold
 Emergency Call
 Alarm Call
 Broadcast/Broadcast group
 Intercom/Multicast
 Call pickup/pickup group
 Call Routing Group, Ring Group
 Call Queue
 Coloring Ring Back Tone(CRBT)
 Custom Prompt, Distinctive Ringtone

Deployment platforms

Hardware Architecture: X86/ARM
Virtual Machine:
 VMware, Fusionsphere, FusionComputer, KVM
Cloud Deployment:
 Alibaba Cloud, Google Cloud, Telecom Cloud, Amazon Cloud, etc.
Network
 QoS, NAT, Fail2ban
 HA hot-standby
Maintenance
 Web GUI Configuration
 Command Line Configuration
 Configuration Restore/Backup
 Multiple Languages Support
 HTTP/TFTP/FTP Firmware Upgrade
 CDR Report and Export
 Ping and Tracert Test
 Network Capture
 NTP
 Internet Quality Test
 Users with Multi-level Permission
 API
 Network Space Storage